

METHOD AND DEVICE FOR PROVIDING CONFERENCES

CLAIM FOR PRIORITY

This application claims priority to German application
5 Application No. 10238286.7 which was filed in the German
language on August 21, 2002, the contents of which is
hereby incorporated by reference.

TECHNICAL FIELD OF THE INVENTION

10 The invention relates to a method and a device for
providing conferences, and in particular, by providing a
control interface between a recognition or
announcement/dialog function and a conference management
function.

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BACKGROUND OF THE INVENTION

With the prior art, services for audio conferences are
part of the essential range of services offered by voice
switching networks. They are provided by the switching
20 centers of the network or even by network elements
external to the switching centers. The conference
function here is based on a combined function for the
audio stream of the participating conferees, which is
provided by a specific hardware unit with DSP capacity
25 (Digital Signal Processor).

In conventional cases, where the useful channel of a
connection is fed to the switching center, the conference
functions and the announcement and tone functions
30 required for these can be provided by peripheral devices
or external devices equipped with corresponding
functionality. If, however, the useful data is conveyed
outside the switching center in a packet network,
preferably at least one external conference system is
35 used for this. The system has interfaces with the packet
network for the useful data of the conference. The useful
data of the conference here is either the useful data of
the individual conferees or the announcements/dialogs and

tones to be input as well as the combined signal to be distributed to the conferees, which is generated via at least one conference bridge. The external conference system can also have a control interface with the 5 switching center controlling connections in the packet network conveyed outside the switching center, in order to control the required basic functions during the conference or to initiate the interspersion of announcements/dialogs and tones generated in the external 10 conference system for example.

Essentially conference services have a range of conference features, which can be differentiated and defined in respect of the initiation and control of the 15 progress of the conference:

There are on the one hand conference features, with which users are included as participants by DIAL-IN (dialing of the conferee into the conference) or by DIAL-OUT (calling 20 of the conferee out of the conference), i.e. the conference process is characterized by the availability of the conferees (e.g. by connecting participants to the conference or the departure of participants from the conference).

25 On the other hand, there are conference features which are characterized by the conference leader or the conferees of the controlled conferences. For example, conferees can be connected, switched to silent or 30 disconnected from the conference by a conference leader by means of appropriate DSS1 signaling (ETSI ADD-ON conference) or via an additional graphic control system on a PC-type terminal. These conference features controlling the conference are often available to the 35 Conference Service Operator, who can manage the conference resources in the network and monitor the conference service.

With regard to videoconferences, which are increasingly used in packet-based networks, the need for conference control is increased by the participating conferees, who increasingly wish to influence the image to be viewed.

5 This includes the selection of one or more participants during the conference, voice-activated switching of the image to the conferees speaking at the time, simultaneous image availability for a certain number of conferees and the additional insertion of documents.

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Conversely, existing conference solutions inform the participants currently in a conference about the inclusion of a further conferee in the conference or the fact that a conferee has left the conference by means of 15 conference tones and/or by means of generally few conference announcements of corresponding content.

With regard to the initiation and control of conferences, the following distinction is made between conference 20 services:

With the ETSI-ADD-ON conference, control is by definition only possible in a local switching center. It is initiated and controlled via conferee signaling 25 (numerical sequence control). It is primarily available in TDM-based but also in packet-based networks, the switching centers of which support conventional participant signaling and can be initiated directly (AD HOC).

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The PRESET conference represents a compromise between AD HOC initiation and a simultaneously predefined conferee list.

35 The PHONEMEET conference is offered as a general network service (public conference). This service, which is very similar to the internet chat service but is much longer established, provides a Service Code, which can be used

to dial into a conference on a specific topic and have discussions with conferees who have already dialed into the topic. Conferees do not generally identify themselves and have no guarantee that they will be connected to a 5 repeat joint conference when they dial in again. The characterizing feature of such a service is that participants, who generally do not know each other, can have discussions in the public network. No control by conferees is required, and automatic monitoring of 10 disruptive parties is not available. Some network operators have operators to monitor conference availability and the undisrupted progress of the conference, the operators identifying and isolating hostile disruptive parties by sporadically listening in. 15

Pre-reserved conferences are available as DIAL-IN, DIAL-OUT or MIXED DIAL-IN/DIAL-OUT conferences. They are particularly useful for business customers. One disadvantage is that pre-reservation and conference 20 planning have to be carried out manually and there is therefore no AD HOC availability.

For the purposes of completeness, reference should be made to conference services with Web-based operator 25 interfaces (such as Siemens SURPASS WEBCONFER) and TERMINAL conferences, which are supported according to certain signaling standards. The former can be booked and controlled via internet access. The advantage of Web-based control with Status Display is limited by the 30 disadvantage of internet access with the possible requirement of an additional terminal for the conference leader and lack of interaction with the conferees. TERMINAL conferences are for example conferences for 35 audio, video and data, which depend on the terminal functions and are possible with the specifications of the H.323 Standard (or even the SIP Standard), with which conventional terminals cannot be used. A central bridge is superfluous here. Major conferences with a large

number of participants are however not possible due to the limited performance of the terminals. A further disadvantage is the increased bandwidth requirement between conferees.

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Resources have to be made available in the network for all conference services. As conference services represent a cost-intensive investment for network operators, they are not made available to an unlimited degree in the 10 network. This means increased control costs; for as well as the interactions between the conference leader and the participants during the conference, the time and date of the conference have to be agreed, the availability of the conferees and the appropriate conference resources has to 15 be established and participants have to be informed of the time and access authorization, to ensure the success of the conference.

Conferences which can be initiated on an AD HOC basis 20 from experience have a control interface characterized by numerical sequence control of the telephony or a graphic control interface connected to a higher quality, intelligent, possibly additional, terminal, which set the limits for the sporadically immediate operability of any 25 terminal. The system tones and announcements made available to the conferees only allow general conclusions about the progress and status of the conference. As far as pre-reserved conferences are concerned, in some circumstances significant manual interaction is required 30 before the start of the conference. Such impediments make the deployment, use and success of conference solutions problematic.

SUMMARY OF THE INVENTION

35 The invention generally relates to simplified operation and control of conference services.

In one embodiment of the invention, there is a control interface provided between an announcement/dialog function and a conference management function, and a voice recognition functionality is provided. This 5 considerably simplifies the preparation and operation/control of conference services. For example, the functional input via numerical sequences controlling the conference is no longer necessary, as the techniques based on the voice recognition functionality support a 10 user-friendlier dialog between man and machine. Input can therefore be easily corrected by participants or operators. Essentially, operation is simplified for all currently known conference types in TDM and packet-based network environments, in particular for DIAL-IN, DIAL- 15 OUT, MIXED DIAL-IN/OUT, ETSI ADD-ON, PHONEMEET with/without operator monitoring, PRESET conference.

The conference functionality, in particular of conferences initiated on an AD HOC basis, is also 20 decoupled from participant signaling and made available at remote level (i.e. more generally in further switching systems possibly belonging to competitor network operators). Additional devices with graphic operator interfaces are not necessary and operator input is 25 minimized in respect of monitoring, booking and/or the procedural organization of conferences.

One advantage of the invention is the positive impact on conference services. The ETSI ADD-ON conference 30 functionality is not only available and easy to operate in the local switching center, but also in any TDM-based or packet-based network through the use of techniques based on (DTMF) voice recognition, to overcome the restrictions of participant signaling. Techniques based 35 on voice recognition are advantageously used, when compressive coding methods are used particularly in packet-based networks, which do not guarantee disruption-free DTMF transmission. Booking and management processes

for conference services can also be automated using corresponding IVR logic (Interactive Voice Response), such as recognition of disruptive parties and intercept initiation and control of follow-up activities by voice 5 recognition mechanisms.

BRIEF DESCRIPTION OF THE DRAWINGS

The invention is described in more detail below with reference to the drawings, in which:

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Figure 1 shows the basic relationships in the network.

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Figure 2 shows the network elements and interfaces incorporated to eliminate disruptive parties in a conference.

DETAILED DESCRIPTION OF THE PREFERRED EMBODIMENTS

Fig. 1 shows the basic relationships according to the invention. In this embodiment, there is a public TDM-based or packet-based, in particular IP-based network, in which at least one announcement and dialog function IVR, one conference management function KM and at least one conference function KF are available in order to provide user-friendly conference services.

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Fig. 1 also shows, for example, 4 users B1 to B4, who wish to take part in a conference. These 4 conferees are served and controlled by a switching system Vst. Fig. 1 also shows a number of mutually independent conference 30 systems K, in which the conference function KF is operating. Interface devices MCU can also be seen to be part of a conference system and these should be seen as the ends of the useful data streams from and to the conferees.

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The conferees' useful data is switched through under the control of the switching center Vst and fed to these

interface devices MCU. The useful data streams are also combined here.

The conference function KF essentially represents a

5 conventional combined function of multiple input signals for audio or video signals. It also supports the distribution function for further information as well. The platform to be provided for this is characterized by telephony interfaces for adaptation to the network and

10 signaling as well as by DSP-based combined functions for the audio stream and where appropriate further functions for controlling video output (e.g. Voice Activated Video Switching).

15 Conference features such as DIAL-IN or DIAL-OUT are supported in conference connections by the conference function KF and their descriptive data is supplied via a control interface to a conference management function KM. The latter can intervene in a controlling manner at any

20 time in the configuration of an ongoing conference via this interface. The SNMP protocol is used for example as the protocol between the conference management function KM and the conference system K.

25 There are conferees who are distinguished in that their input useful data stream is accessed before inclusion in the combined conference signal and fed to an announcement and dialog function IVR for a certain time for the purposes of monitoring for disruptive activity or to

30 identify legitimate input controlling the conference (e.g. by the conference leader). An announcement and dialog function IVR can be permanently or temporarily assigned here via a control interface S between the announcement and dialog function IVR and the conference

35 management function KM.

The announcement and dialog function IVR operates on at least one separate device or if necessary even collocated

with the function KM described in more detail below on a device VoxP. It is used for dialog management with input recognition for the conference leader or the conferencees, with DTMF input, menu-driven dialog or preferably keyword spotting in the natural dialog being used. The hardware platform required for the announcement and dialog function IVR is generally characterized due to the performance required in public networks by telephony interfaces, which undertake adaptation to the network technology and signaling, as well as by hardware and software, which carry out voice recognition tasks (e.g. DSPs, voice recognition algorithms).

The conference-specific dialog processes necessary for the announcement and dialog function IVR are stored appropriately on a content server CS, e.g. in the form of VoiceXML scripts, which are produced based on the conference configuration and give the complete dialog sequence for the IVR system.

Fig. 1 also shows a conference management function KM, which is configured as a software function and which operates on a device VoxP. This monitors and supports the status of the conference systems K and their ports generally and where necessary network-wide. A further functionality is the reservation of conferences booked in advance, the prompt activation and monitoring/control of the conferences themselves and the generation of charge tickets, in particular with regard to the reservation of resources in the network. Booking data and charge data and where necessary error indices as well as traffic and statistical data are stored on a database server DB by the conference management function KM.

According to one embodiment of the invention, the announcement and dialog function IVR has a control interface S with the conference management function KM, with which it is able to output booking data for a

conference or the initial conference parameters of an AD HOC conference to the conference management function for further processing. Conversely, the announcement and dialog function IVR receives information where necessary

5 about the resource requirement, which can be covered at present or for the intended booking period from the conference management function and where necessary charge information for configuring the dialog with the party ordering the conference/the conference leader.

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The conference management function has an overview of the availability of conference resources, if necessary network-wide, and can therefore in particular support and reserve conferences which extend over a number of

15 conference systems (cascading) because of their size or in the event of resource shortages.

Finally, for reasons of fail-safety, the database server DB, content server CS, conference management function KM, 20 announcement and dialog function IVR and the conference function KF are at least duplicated. The functions do not necessarily have to be set up on different hardware platforms. The IVR function and KM function in particular can be set up on different hardware platforms.

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Fig. 2 shows an application of the method according to the invention, where conferees emitting disruptive signals, hereafter referred to as disruptive parties, can be eliminated in an automated manner from a conference.

30 This means that an automated operator operates in the place of a human operator. A PHONEMEET conference can for example be considered here. As this conference service is subject to charges, the network operator is concerned to ensure and monitor the availability of the service.

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Automated operators are therefore preferably used. These preferably connect in succession to an ongoing PHONEMEET conference, to verify that no disruptive parties are

participating in the conference. The latter are typically characterized by the inputting of disruptive tones or by abusive language. Conferees disrupting in this way are then eliminated by the operator from the conference or at 5 least switched to silent. Attempts will also be made to proceed in the same manner in the event of indecent contributions or even legally prohibited content.

In order that the operator can eliminate just the 10 disruptive parties from the conference when such are identified, without having to disconnect the entire conference, the control unit of the conference system allows the operator to dial in or switch to the entire conference and to dial in or switch to each conferee in a 15 PHONEMEET conference. This means that the operator is able to select the conference in question and thus locate a possible disruptive party.

Fig. 2 shows the network elements and interfaces required 20 for this. The processes are preferably automated using voice recognition algorithms with key word spotting. In this way a recognition system or even an IVR system obtains access to PHONEMEET conferences in succession and monitors the conference (e.g. K1, K2, K3) first. If a 25 disruptive party is identified on the basis of high sound levels, identification of abusive language by the voice recognition system or other criteria, the recognition system/IVR system switches in succession to each individual conferee, to carry out a detailed check. The 30 recognition system/IVR system obtains the data for switching via the conference management system from the database DB, which includes the number of PHONEMEET conferences and their characteristics, such as maximum number of participants and accessibility for the monitor 35 function. Switching can be achieved by dialing into the conference system K, as shown in Fig. 2 or via the control interface of the conference system. The conference system identifies from the signaling of the

higher order switching center, e.g. from the A call number of the recognition system/IVR system or access via the control interface that this is a routine switch on the part of an operator function, which does not generate 5 perceptible indices in respect of the conferees being checked or other conferees.

If a switch to a specific conferee is requested, the conferee's input stream into the conference is also 10 forwarded to the recognition system/IVR system. This is shown symbolically in Fig. 2 by the insertion of a further intercept point, which in some circumstances does not however reflect the actual hardware relationships. The switch to one conferee or the entire conference can 15 be terminated via a command interface or by the recognition system/IVR system or the conference management system canceling the switch connection.

When an individual disruptive party is detected and 20 identified, the recognition system/IVR system prompts the conference system to isolate it. This can take place preferably via the control interface of the conference system but can alternatively be achieved by appropriate dialing in to signal the wish to disconnect or silence a 25 specific conferee. If the disruptive party is switched to silent, it can continue to listen to or watch the conference or alternatively an announcement can be played by the IVR system to clarify isolation status and the follow-up measures taken (switching to silent or 30 disconnection from the conference). Expediently, the conference management system stores the data of the disruptive party (e.g. the A call number), as available, in a data storage unit, e.g. in the database of the content server CS.

35 The prompt by the monitor function to eliminate the disruptive party can be given via the operator interface of the VoxP device. Monitoring details are stored in the

form of VoiceXML scripts on the content server CS. The VoiceXML pages including the conference points currently available and to be checked can be generated by the content server CS using the status parameters obtained 5 from the database DB and supplied by the conference management function KM.

As an alternative for small network configurations with, for example, one conference server, the 10 recognition/IVR/conference management function can also operate on the conference server itself.

In the above application the IVR system is used as a recognition system. If clarifying announcements are to be 15 played to the isolated participants, a recognition system with announcement functionality is preferably used. The availability of a full interactive recognition and response function, in other words full IVR functionality, is required if the disruptive party, when switched to 20 silent, is to be offered a dialog offering a range of different options (e.g. complaint function, etc.).